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Case No. 8285/167

## PATENT APPLICATION TRANSMITTAL LETTER

To the Assistant Commissioner for Patents:

Transmitted herewith for filing is the patent application of: Robert F. Manning et al. for : SYSTEM AND METHOD FOR ROUTING BOTH TOLL-FREE AND CALLER-PAID TELEPHONE CALLS TO CALL SERVICE CENTERS. Enclosed are:

- ☒ 4 sheet(s) of drawings, 16 pages of application (including title page), and the following Appendices : \_\_\_\_.
- ☐ Declaration.
- ☐ Power of Attorney.
- ☐ Verified statement to establish small entity status under 37 CFR §§ 1.9 and 1.27.
- ☐ Assignment transmittal letter and Assignment of the invention to : \_\_\_\_.
- ☐ \_\_\_\_.

Claims as Filed	Col. 1	Col. 2
For	No. Filed	No. Extra
Basic Fee		
Total Claims	-20	
Indep. Claims	-3	
Multiple Dependent Claims Present		

\*If the difference in col. 1 is less than zero, enter "0" in col. 2.

Small Entity	
Rate	Fee
	\$ 395
x\$11=	\$
x\$41=	\$
+\$135=	\$
Total	\$

Other Than Small Entity	
Rate	Fee
	\$ 790
x\$22=	\$
x\$82=	\$
+\$270=	\$
Total	\$

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- ☐ The issue fee set in 37 CFR § 1.18 at or before mailing of the Notice of Allowance, pursuant to 37 CFR § 1.311(b).

June 12, 1998  
Date

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Registration No. 37,834

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PATENT  
Case No. 8285/167

IN THE UNITED STATES PATENT AND TRADEMARK OFFICE  
APPLICATION FOR UNITED STATES LETTERS PATENT

INVENTORS:

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TITLE:

SYSTEM AND METHOD  
FOR ROUTING BOTH  
TOLL-FREE AND  
CALLER-PAID  
TELEPHONE CALLS TO  
CALL SERVICE CENTERS

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# SYSTEM AND METHOD FOR ROUTING BOTH TOLL-FREE AND CALLER-PAID TELEPHONE CALLS TO CALL SERVICE CENTERS

## BACKGROUND OF THE INVENTION

The present invention relates to managing telephone calls directed to call  
5 service centers. More particularly, the present invention relates to a system and  
method for routing both toll-free and caller-paid telephone calls to call servicing  
centers.

Many business enterprises make use of toll-free numbers, such as 800, 877  
or 888 numbers, to provide customers with call services that allow customers to dial  
10 in and speak to company representatives or hear a recording. The call service is  
supported by one or more call service centers that may handle such functions as  
product ordering, dissemination of product information, or general company  
information. Customers typically expect prompt attention to their inquiries and thus  
may avoid doing business with companies that use telephone services which place  
15 customers on hold for extended periods of time. Both the customer and the  
company providing the call service desire efficient call distribution and handling.

If a company has a call service that uses a single call center with a group of  
agents available at one location, the company provides its customers with a toll-free  
number that is routed to an automatic call distribution device(ACD). The ACD is  
20 linked to the various agents at the call service center and can route calls based on  
agent availability and expertise. While this method of call distribution is effective for  
handling telephone calls directed to a single call center having agents in the same  
geographical area, it does not address the need for handling calls over a toll-free  
line to a business enterprise having multiple call centers in various geographical  
25 locations.

One solution to managing multiple call service centers that are geographically  
dispersed is provided by the GEOTEL Intelligent CallRouter® (ICR). The ICR  
service provides interfaces to multiple carrier networks and provides call routing to  
agents at call service centers independent of the particular toll-free network provider  
30 used. Present telephone systems utilize the ICR to manage incoming call center

traffic over toll-free lines and do not address caller-paid calls (*i.e.*, calls billed to the caller). Business enterprises that offer call services may wish to use a local telephone number to maintain a local presence to their customers even though the local number may connect to a call service center in another geographical location.

5 Additionally, toll-free numbers can be a large expense for a business enterprise and some businesses may choose to convert more of their telephone traffic to regular telephone numbers if the local (caller-paid) numbers can offer the same level of service to their calling customers.

Accordingly, there is a need for an improved system and method for  
10 managing call distribution to call service centers that can handle both caller-paid and toll-free telephone calls.

#### BRIEF DESCRIPTION OF THE DRAWINGS

FIG. 1 is an illustration of a system for managing caller-paid and toll-free telephone calls according to one embodiment of the invention;

15 FIG. 2 is a block diagram of an advanced intelligent network for use in the system of FIG. 1;

FIG. 3 is a block diagram of an intelligent processing platform interface;

FIG. 4 is a flow diagram showing a preferred method of processing a caller-paid telephone call using the system of FIG. 1;

20 FIG. 5 is a flow diagram of an alternative method of processing a caller-paid telephone call in an advanced intelligent network;

FIG. 6 is a flow diagram of a third alternative embodiment of a method of processing a caller-paid telephone call in an advanced intelligent network.

#### 25 DETAILED DESCRIPTION OF THE PRESENTLY PREFERRED EMBODIMENTS

Figure 1 illustrates an embodiment of a system for routing both toll-free and caller-paid telephone calls according to the present invention. The system 10 includes one or more interexchange networks 12, a local exchange network 14 and an intelligent processing platform (IPP) 16. The interexchange networks 12 may be  
30 any one of a number of telephone systems that handle toll-free telephone calls, typically 800, 877 or 888 telephone calls, and handle transport of telephone calls

across local access transport area (LATA) boundaries. The interexchange networks 12 communicate with the IPP 16 via data channels 18. The interexchange networks 12 carry telephone voice and data information over voice channels 20. The local exchange network 14 also communicates with the IPP 16 via a data channel 22.

- 5 Voice and information from a caller utilizing a local exchange network 14 is transmitted over voice lines 24.

The system 10 also includes a call service having at least two call service centers 26 positioned in different geographical locations. Each call service center 26 includes a group of agents 28. An automatic call distributor (ACD) 30 receives data  
10 information regarding the voice lines 20, 24 utilized by the local exchange network 14 or interexchange network 12. The ACD 30 also routes the incoming call to the appropriate agent 28. The ACD 30 keeps track of, among other things, the status of agents and the estimated wait time for reaching an agent. Any of a number of commonly available ACDs may be used. Examples of suitable ACDs 30 are the  
15 Lucent Technologies Definity ACD and the Meridian from Northern Telecom Limited. In one alternative embodiment, a local central office switch such as a Lucent Technologies 5ESS may provide the services of a typical ACD. The ACD 30 may be a stand-alone device or integrated with a private branch exchange (PBX).

Each ACD 30 is preferably connected to a gateway 32 via a datalink 34. The  
20 gateway 32 is also linked via a data line 36 to the IPP 16. Each gateway 32 contains software specific to the type of ACD 30 capable of translating agent status information maintained at the ACD 30 into a format understandable by the IPP 16. In this manner, various types of ACDs 30 may be used in the system 10. The IPP 16 may be a single processor or a distributed network maintaining a log of agent 28  
25 status at the various geographical locations based on information provided from the ACDs 30 to the gateway 32. The IPP 16 processes the information from the gateways 32 to provide a centralized control of call distribution from the interexchange networks 12 and local exchange network 14. Suitable devices for the IPP may be a GeoTel Communications ICR or a Genesys Telecommunication  
30 Laboratories Intelligent Call Distributor.

FIG. 2 illustrates a preferred local exchange network 14 having advanced intelligent network (AIN) capabilities. The local exchange network 14 includes one

or more end office switches equipped as AIN service switching points (SSP) 38. The SSP 38 is a programmable switch having the ability to recognize AIN triggers for calls requiring special services. The SSP 38 may be an end office or tandem switch and communicates with a service control point (SCP) 40. An end user may be  
5 connected to the SSP 38 over a voice/information channel 24 such as an ordinary telephone line. Multiple connections and combinations of network elements are usable in the local exchange network 14. For example, an end user may also connect to an SSP 38 through one or more ordinary telephone switches.

The SCP 40 is a network element in the AIN local exchange network 14  
10 containing logic and data necessary to provide the functionality required for the execution of a desired communication service. An SCP 40 generally permits separation of service logic from switching functionality such that additional services may be developed without the need to provision software in each individual SSP 38. A suitable SCP 40 is the Advantage SCP manufactured by Lucent Technologies.  
15 The SCP 40 is preferably in communication with the SSP 38 via a signal transfer point (STP) 42. The STP 42 routes signals between different network elements. A suitable data signaling standard for use with the STP is the American National Standards Institute (ANSI) signaling system number 7 (SS7). The SCP 40 preferably communicates with the IPP 16 over a data line 22. The SSP 38 in the  
20 local exchange network 14 may communicate with an intelligent peripheral 44 over a data 46 or voice 48 channel.

The IP 44 is a network element of the AIN that contains resources to exchange information with an end user and perform other functions such as call origination and tone generation. The IP 44 provides special resources for  
25 interactions between the end user and the network such as dual tone multi-frequency (DTMF) recognition, playing pre-recorded announcements and tone generation. A service node/intelligent peripheral (SN/IP) platform manufactured by Comverse Technology, Inc. is suitable for use with the AIN local exchange network 14. Although the local exchange network 14 illustrated in FIG. 2 shows only one of  
30 each network element, those of ordinary skill in the art understand that the presently preferred system and method may include more complex networks having a plurality of interconnected SCPs, IPs, STPs, and SSPs.

Referring to FIG. 2, the SCP 40 preferably includes an IPP interface 50 for generating messages to, and interpreting messages from, the IPP 16. As shown in FIG. 3, the IPP interface 50 generates a routing query message 52 and expects a routing response message 54 in return. The routing query message includes a message header 56 and a transaction ID 58 identifying the type of message and a unique number assigned to the telephone call, respectively. The routing query message 52 also includes the dialed number 60 and the calling party's number 62. The routing query message 52 may optionally include a segment containing collected digits 64 so that any digits collected through a play and collect menu in the local exchange network 14 may be passed on through the agent 28 at the call service center 26 assigned to the telephone call. Another optional field in the routing query message is an optional parameter field 66 which may be used for future expansion and adaptability of the network 10. The routing response message 54 also includes a header 68 and transaction ID 70 along with the route response 72 generated by the IPP 16. Additionally modified or new parameters 74 are optionally included in the message returned by the IPP 16. The route response 72 may be a ten digit number or a trunk group identifier with outpulse digits. The IPP interface 50 also preferably includes default instruction logic 76 for routing a caller-paid telephone call originating in the local exchange network to one of the service centers 26 in the event that there is no response from the IPP 16. In one embodiment, the interface between the SCP 40 and the IPP is TCP/IP. Other communication interfaces, such as SS7 may also be used.

Utilizing the network 10 described above, a preferred method of utilizing the common architecture for routing both caller-paid telephone calls from a local exchange network 14 and toll free telephone calls from interexchange networks 12 using an intelligent processing platform 16 is described below. Referring to FIG. 4, a caller-paid telephone call in a local exchange network 14 comes in over a voice line or trunk to the SSP 38 (at step 76). The telephone call then encounters an AIN trigger (at step 78). Recognizing that the telephone call requires AIN processing, the SSP sends a query to the SCP 40 via the STP 42 for routing instructions (at step 80). The service logic 51 in the SCP 50 recognizes the call is directed to a call service center and will require the generation of a routing query message 52 to

query the IPP for routing instructions (at step 82). The SCP next launches the query to the IPP 16 (at step 84). The routing query message 52 includes at least the dialed number 60. The message may also contain the calling party's number 62 and the other information described above.

5 In an alternative embodiment, the routing query message sent from the SCP to the IPP may only include a portion of the calling party's number 62 in order to preserve privacy. For example, if the local exchange network containing the SCP is owned by a first enterprise and the IPP is owned by a second enterprise, the first enterprise may only wish to disclose a portion of the calling party's number to the  
10 IPP. Preferably, the routing query message 52 contains enough of the calling party's number, such as the area code or the area code and prefix, to allow the IPP to make an informed routing determination.

In response to the routing query message the IPP accesses the customer routing script 63 associated with the dialed number 60. The customer routing script  
15 63 is essentially a flow-chart or a list of rules indicating the actions to be taken when the particular dialed number 60 is received. Using the script 63 and information about the real-time activities at the customer's call centers 26 gathered by the gateways 32, the IPP 16 makes a routing determination (at step 86). The IPP then returns the routing destination in the routing response message 54 and sends it to  
20 the SCP 40 (at step 88). The route response 72 may be a 10 digit number or a trunk group with outpulse digits. Alternatively, the route response may be an international telephone number. The SCP 40 communicates with the SSP 38 over a data line connected by the STP communicating the routing destination for a telephone call (at step 90). The SSP subsequently routes the call to the indicated routing destination  
25 (at step 92). Separately or concurrently with the one or more telephone calls directed to the IPP from the local exchange network 14, the interexchange networks 12 may also be querying the IPP 16. The IPP 16 preferably gathers all the call information generated at the ACDs 30 and transmitted via the gateways 32 to make call routing determinations to optimize the use of agents 28 at the various call  
30 service centers 26.

In instances where there is a failure of communication between the SCP 40 and IPP 16, the SCP may utilize default instruction logic 76 to determine when



default routing should be used. For example, in one embodiment, the SCP may wait for a predetermined interval before accessing the default instructions and then will route the call to a single predetermined number. Alternatively, the default instructions may include directions to send calls to one of several predetermined numbers on a percent allocation basis. Thus, the SCP would instruct the SSP to direct the call to one of a number of predetermined default call center destinations.

As shown in FIG. 5, an alternative method of processing a caller-paid telephone call may include the SSP playing a message and collecting digits of the caller's response to the message. As in FIG. 4, when a call comes in to the SSP and it is recognized as an AIN call and as one destined for a call service center (FIG. 4, at steps 76-82). In this example, the service logic 51 indicates that a message needs to be played to the caller (at step 94). The SCP 40 instructs the SSP to play a particular announcement and collect digits from the caller (at step 96). This step may be repeated several times depending on the instructions from the SCP 44. In response to the SCP's instructions, the SSP plays the announcement, collects the digits from the caller and passes on the caller-entered digits to the SCP (at step 98). The SCP 40 then launches a query to the IPP including at least the dialed number 60 and the collected digits 64 (at step 100). The IPP 16 then accesses the customer routing script 63 associated with the dialed number, and then based on the routing script, the caller-entered digits and the information on the real time activities at the call centers, the IPP makes a routing determination at step 102. As discussed earlier, the IPP returns the destination and the call is routed to the appropriate destination (at steps 88-92).

FIG. 6 illustrates an example where an external resource other than an SSP is used interact with the call. As before, the call is first recognized as an AIN call (FIG. 4, at steps 76-82). In this case, however, the service logic 51 indicates that a menu announcement should be played to the caller using an external source such as an IP 44 (at step 102). The SCP 40 then sends instructions to the IP over the data link 46 (via the SSP) indicating that a particular announcement or announcements should be played and digits collected from the caller (at step 104). The IP plays the announcements, collects the digits, and then passes the caller-entered digits to the SCP (at step 106). The SCP 40 then launches a query to

the IPP including at least the dialed number 60 and the collected digits 64 (at step 108). Subsequently, the IPP 16 accesses the customer routing script 63 associated with the dialed number and returns a routing destination to the SCP in a routing response message 54. The SSP then routes the call to the appropriate destination.

5           The present system and method provide flexibility and advantages over the prior art. Business enterprises that are customers of local exchange networks may utilize the present common architecture for directing both local calls and long-distance calls through the common control of the IPP to reduce costs and maintain a local presence while still retaining the advantage of a centralized call distribution network. Thus not only can the same architecture be used, but the calls may be directed through common routing scripts.

Alternative embodiments include connecting to an interexchange network number rather than an interexchange network SCP so that the interexchange network toll-free number will route to a regular local number and then use the local exchange network SCP in a manner similar to that described previously. The caller interaction with a play and collect menu that occurs in the local exchange network prior to the SCP querying the IPP for routing instructions can offer a time and cost savings to the business enterprise supporting the customer call service centers. According to an aspect of the present invention, the SCP for the local exchange network can pass a presentation restriction indicator (*i.e.*, a caller ID block) and other information such as the originating station type (*e.g.*, a pay phone, hotel/motel, etc.). With this information, the routing scripts in the IPP can treat calls differently if desired. Also, the AIN trigger may be a 3/6/10 digit public office dialing plan trigger, a specific digit string trigger, or a termination attempt trigger. The 3/6/10 digit public office dialing plan refers to triggering on a 3 digit number (area code), a 6 digit number (area code + prefix), or an entire 10 digit telephone number.

The system described above may be programmed to allow the AIN service logic 51 to only query for routing instructions in certain cases and not others. For example, the service logic 51 may specify that the SCP query for fifty percent of the calls and route the other calls to the dialed number. In other embodiments, the SCP 40 may query the IPP for further routing instructions in the event that an initial routing instruction given by the IPP results in a busy signal or no answer. In this

case, the SCP will indicate to the IPP that a busy signal or no answer was encountered.

It is intended that the foregoing detailed description be regarded as illustrative rather than limiting and that it be understood that the following claims, including all

5 equivalents, are intended to define the scope of this invention.

WE CLAIM:

1. A system for routing both toll-free and caller-paid telephone calls comprising:

a call service having at least two call service centers;

5 an interexchange network for handling toll-free telephone calls directed to the call service;

a local exchange network for handling caller-paid telephone calls directed to the call service; and

10 a call routing processor in communication with the call service centers, the interexchange network, and the local exchange network, wherein the call routing processor is configured to receive status messages from the call service centers and provide routing instructions to both the interexchange network and the local exchange network, and wherein the toll-free and caller-paid telephone calls to the call service originating at the interexchange network and the local exchange network are routed to an appropriate service center.

2. The system of claim 1, wherein the local exchange network comprises an advanced intelligent network.

3. The system of claim 1, wherein the call routing processor further comprises common call routing scripts for routing telephone calls received from the interexchange network and the local exchange network.

4. The system of claim 2, wherein the advanced intelligent network comprises at least one service control point (SCP) in communication with the call routing processor over a data channel.

5. The system of claim 4, wherein the SCP further comprises a call routing processor interface in communication with the call routing processor over the data channel.

6. The system of claim 5 further comprising an intelligent peripheral in communication with a switch via a data channel and a voice/information channel.

7. A system for routing both toll-free and caller-paid telephone calls, the system comprising:

a call service having a plurality of call service centers

an interexchange network for handling toll-free telephone calls directed

5 to the call service;

a local exchange network for handling caller-paid telephone calls

directed to the call service, the local exchange network having a call controller, and

a call routing processor in communication with the call service centers,

the interexchange network, and the local exchange network, the call routing

10 processor configured to receive status messages from the call service centers and

provide routing instructions to the interexchange network and the local exchange

network, wherein the call controller comprises a call routing processor interface

having communication logic for communicating with the call routing processor.

15 8. The system of claim 7, wherein the call routing processor interface further comprises default instruction logic for routing a caller-paid telephone call to a call service center in an absence of routing instructions from the call routing processor.

20 9. The system of claim 7, wherein the call routing processor interface further comprises service logic responsive to caller-paid telephone calls directed to a particular call service and capable of performing predetermined functions on the caller-paid telephone call.

25 10. The system of claim 7, wherein the communication logic for communicating with the call routing processor comprises a routing query message format, the routing query message format having information on a telephone number of a calling party of a caller-paid telephone call.

30 11. The system of claim 10, wherein the information on the telephone number of the calling party comprises an area code of the calling party, wherein the telephone number of the calling party is not shown to a called party.

12. The system of claim 10, wherein the information on the telephone number of the calling party comprises an area code and a prefix of the telephone number of the calling party, wherein the telephone number of the calling party is not shown to a called party.

13. The system of claim 10, wherein the information on the telephone number of the calling party comprises a presentation restriction indicator, wherein the telephone number of the calling party is not shown to a called party.

14. The system of claim 10, wherein the information on the telephone number of the calling party comprises originating station data, whereby a type of originating station for a caller-paid telephone call is identified.

15. The system of claim 7, wherein the interexchange network comprises a plurality of interexchange networks and the local exchange network comprises a plurality of local exchange networks.

16. A method of routing caller-paid and toll-free telephone calls directed to a call service in system for routing both toll-free and caller-paid telephone calls, the system comprising a call service having a plurality of call service centers, an interexchange network for handling toll-free telephone calls directed to the call service, a local exchange network for handling caller-paid telephone calls directed to the call service, the local exchange network having a call controller, and a call routing processor in communication with the call service centers, the interexchange network, and the local exchange network, the call routing processor configured to receive status messages from the call service centers and provide routing instructions to the interexchange network and the local exchange network, the method comprising the steps of:

receiving a caller-paid telephone call at a switch in the local exchange network;

identifying that the caller-paid call is a call directed to a call service;

requesting call routing instructions from the call routing processor; and

routing the caller-paid telephone call according to the call routing instructions received from the call routing processor.

17. The method of claim 16, wherein the step of requesting call routing instructions comprises examining service logic in a call routing processor interface for instructions on handling telephone calls to a particular call service, executing the instructions and sending a routing query message to the call routing processor.

18. The method of claim 17, wherein the step of executing the instructions comprises the steps of playing a message to a caller and collecting at least one digit entered by the caller in response to the message, and wherein the step of sending a routing query message comprises sending the at least one collected digit to the call routing processor.

19. The method of claim 17, wherein the local exchange network further comprises a switch in communication with the call controller, and wherein the step of executing the instructions comprises the steps of the call controller instructing the switch to play a message and collect at least one digit entered by a user in response to the message, and the switch communicating the at least one digit to the call controller.

20. The method of claim 17, wherein the local exchange network further comprises a switch in communication with the call controller, and an intelligent peripheral in communication with the switch, wherein the step of executing the instructions comprises the steps of the call controller instructing the intelligent peripheral to play a message and collect at least one digit entered by a user in response to the message, and the intelligent peripheral communicating the at least one digit to the call controller.

21. The method of claim 16, further comprising the step of the call controller requesting subsequent call routing instructions from the call routing processor if the routed caller-paid call encounters a busy signal.

22. The method of claim 16, further comprising the step of the call controller requesting subsequent call routing instructions from the call routing processor if the routed caller-paid call encounters a no answer condition.



## ABSTRACT OF THE DISCLOSURE

A system and method of routing both caller-paid and toll-free telephone calls includes a local exchange network and an interexchange network in communication with a call routing processor. The call routing processor receives status information on at least two call service centers and provides routing instructions to both the interexchange and local exchange networks. The method includes the steps of a call controller in a local exchange network communicating with a call routing processor also in communication with an interexchange network, the call controller identifying calls to call service centers and requesting instructions from the call routing processor.

FIG. 1

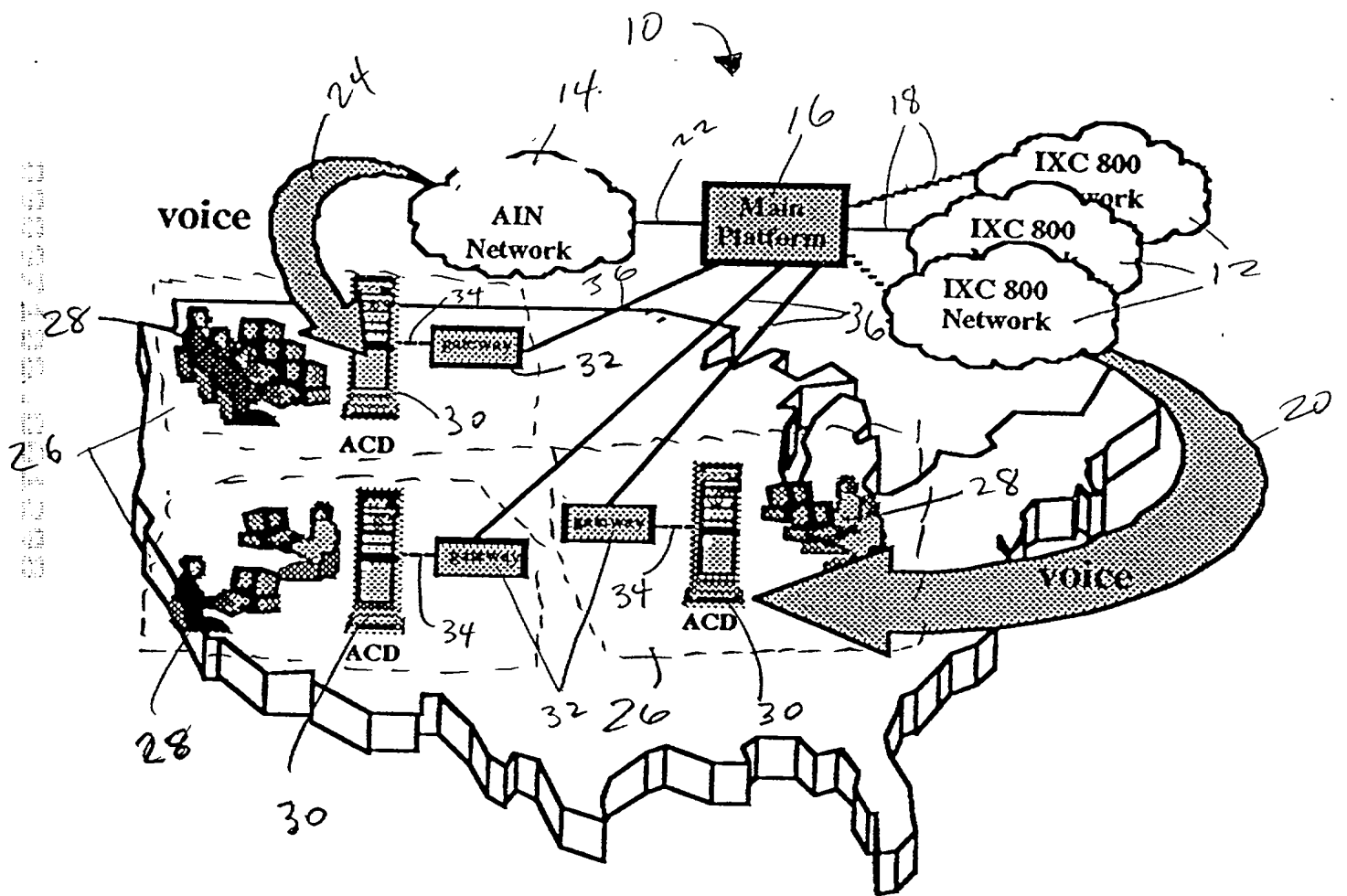


FIG. 2

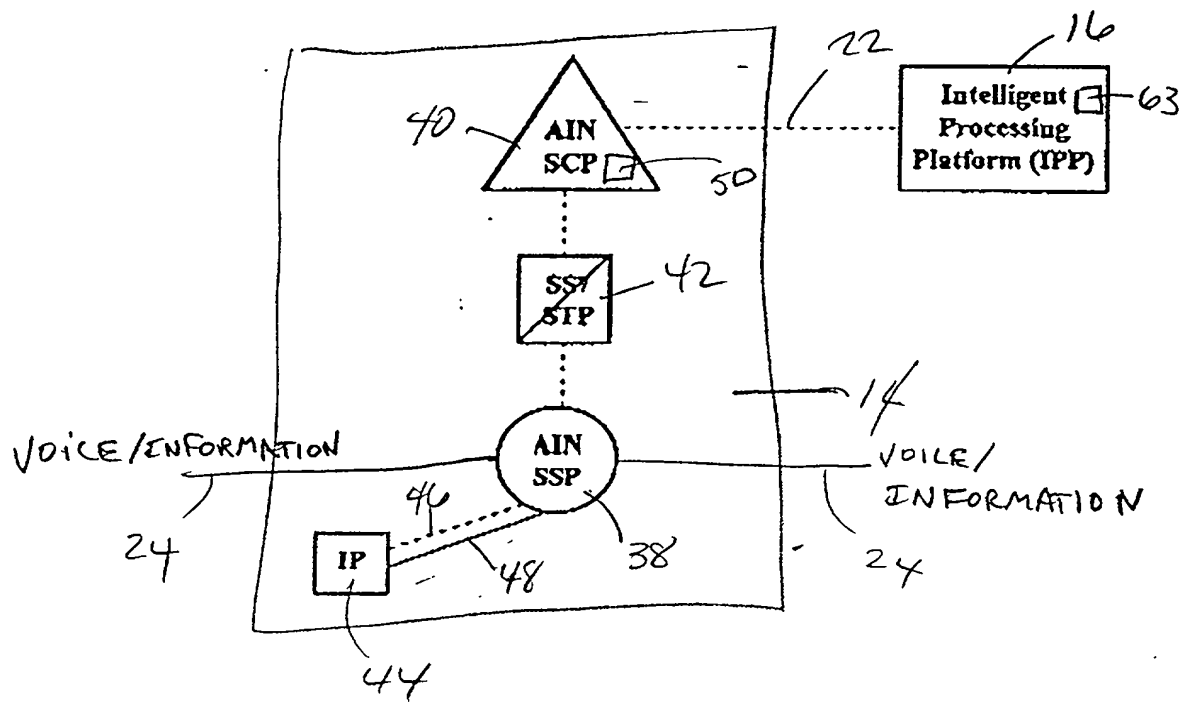
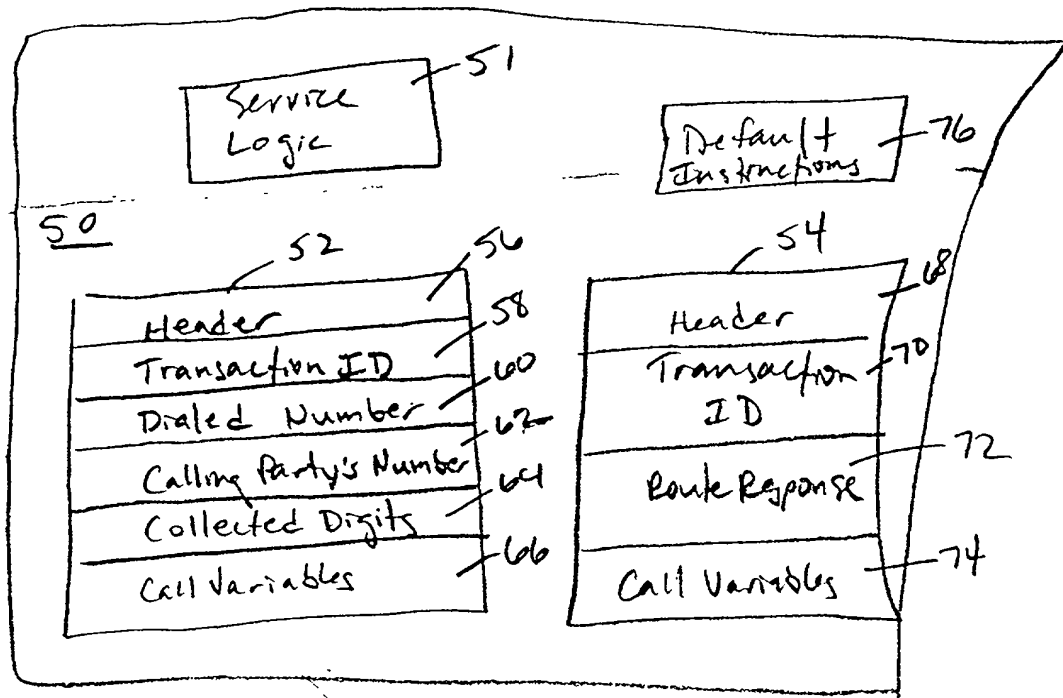
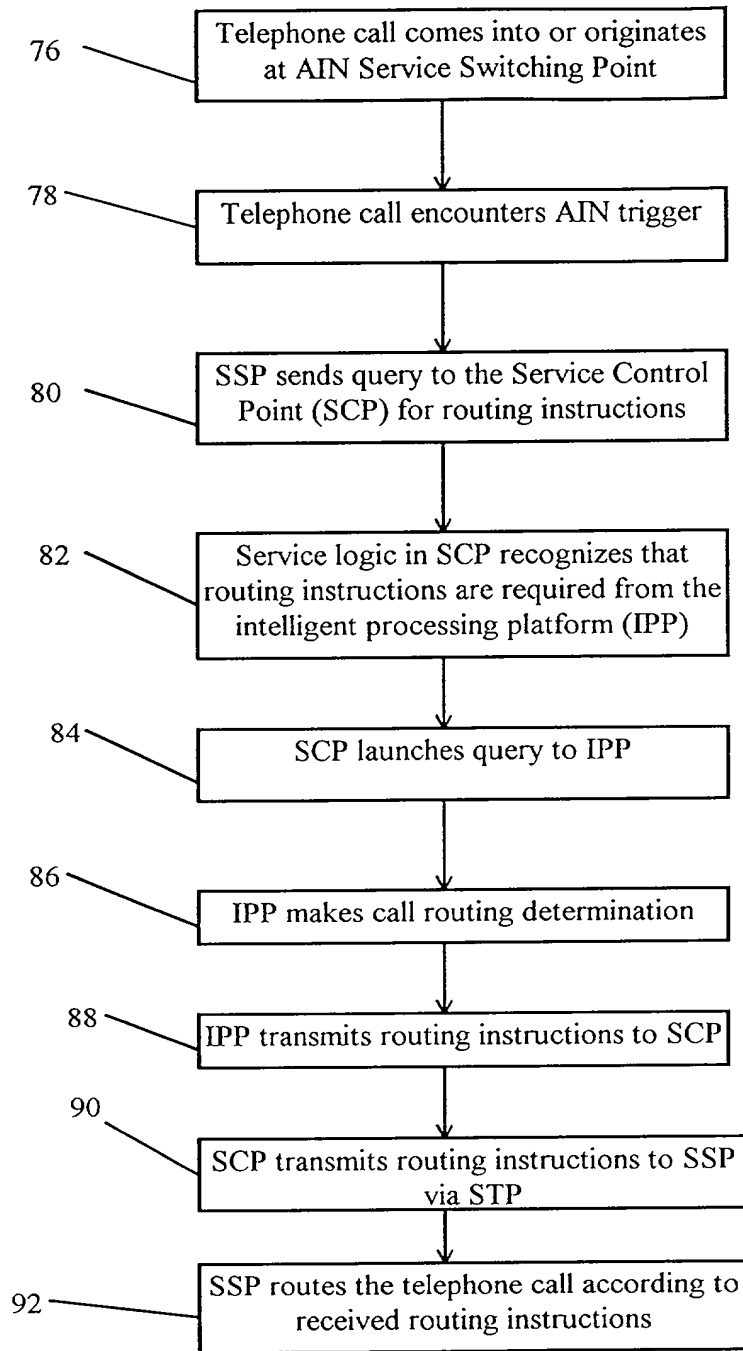


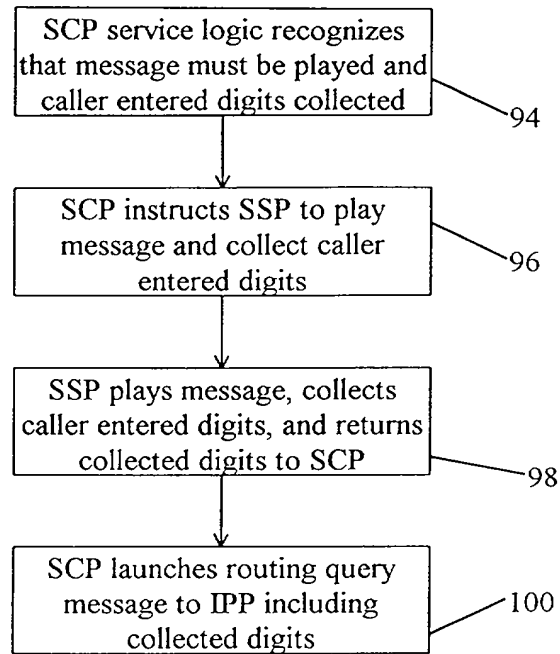
FIG. 3



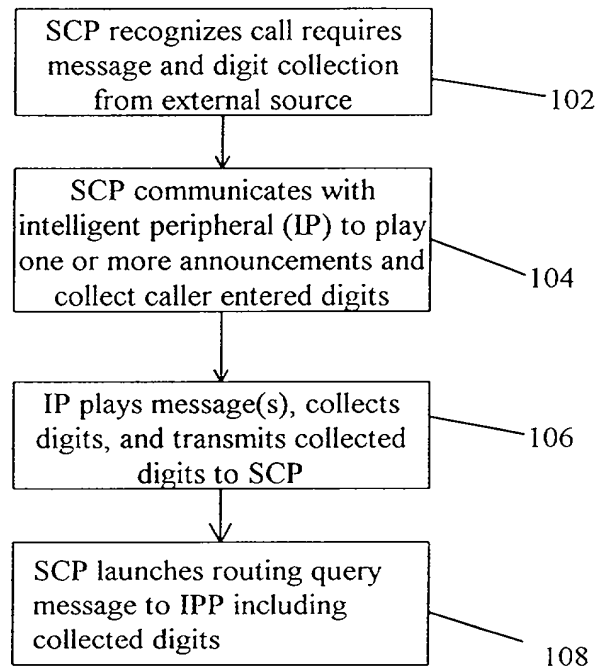
**FIG. 4**



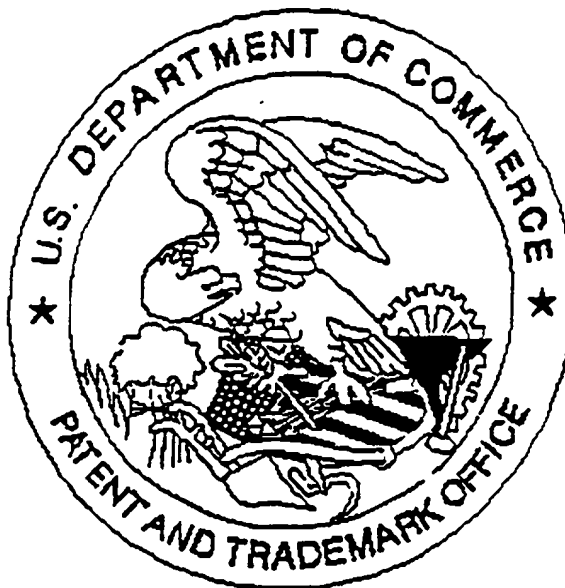
**FIG. 5**



**FIG. 6**



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